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United States Patent [19]**Dent**[11] **Patent Number:** **5,539,730**[45] **Date of Patent:** **Jul. 23, 1996**[54] **TDMA/FDMA/CDMA HYBRID RADIO ACCESS METHODS**[75] **Inventor:** **Paul W. Dent, Stehag, Sweden**[73] **Assignee:** **Ericsson GE Mobile Communications Inc., Research Triangle Park, N.C.**[21] **Appl. No.:** **179,954**[22] **Filed:** **Jan. 11, 1994**[51] **Int. Cl.:** **H04J 4/00**[52] **U.S. Cl.:** **370/29; 370/50; 370/95.1**[58] **Field of Search:** **370/18, 50, 69.1, 370/70, 75, 76, 24, 29, 30, 49.5, 95.1, 95.3, 104.1; 379/60; 455/13.1, 13.2, 33.1, 63, 82, 83**[56] **References Cited****U.S. PATENT DOCUMENTS**

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Primary Examiner—Melvin Marcelo**Attorney, Agent, or Firm**—Burns, Doane, Swecker & Mathis[57] **ABSTRACT**

A radio access method for facilitating communication between at least one first station and a plurality of second stations is disclosed. First, each signal intended for transmission is buffered at a first station. The signals are then divided into equal length segments. The signal segments intended for a particular one of the second stations is transmitted using a corresponding time slot in a regularly repeating time multiplex frame. The signal segments transmitted by the first station are received at at least one of the second stations and the signal segments are assembled from successive corresponding time slots to reconstruct said intended signal. A transmit frequency channel uniquely associated with the corresponding receive time slot is determined at the second station. Finally, a signal intended for transmission to the first station is buffered in the second station and compressed for transmission using the transmit frequency channel during substantially the entire time period that the second station is not receiving.

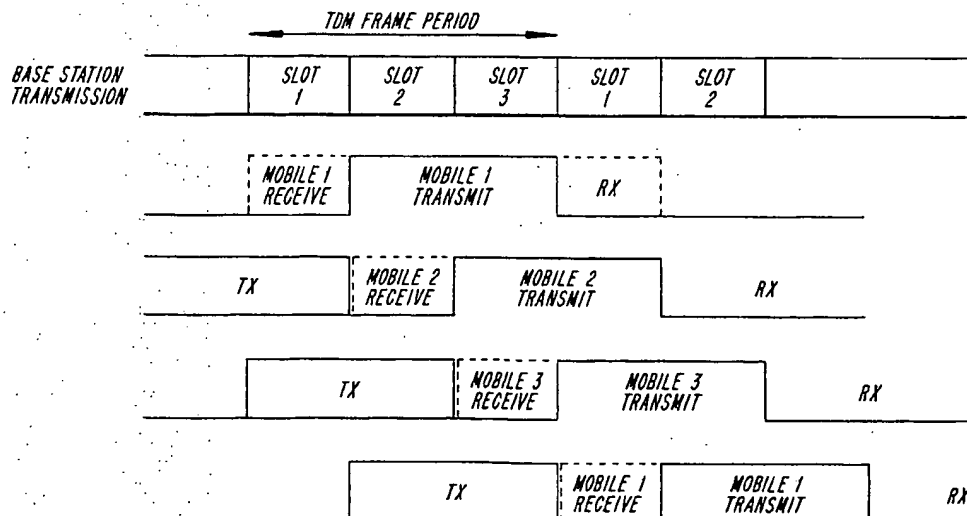
49 Claims, 11 Drawing Sheets

Fig. 1
(PRIOR ART)

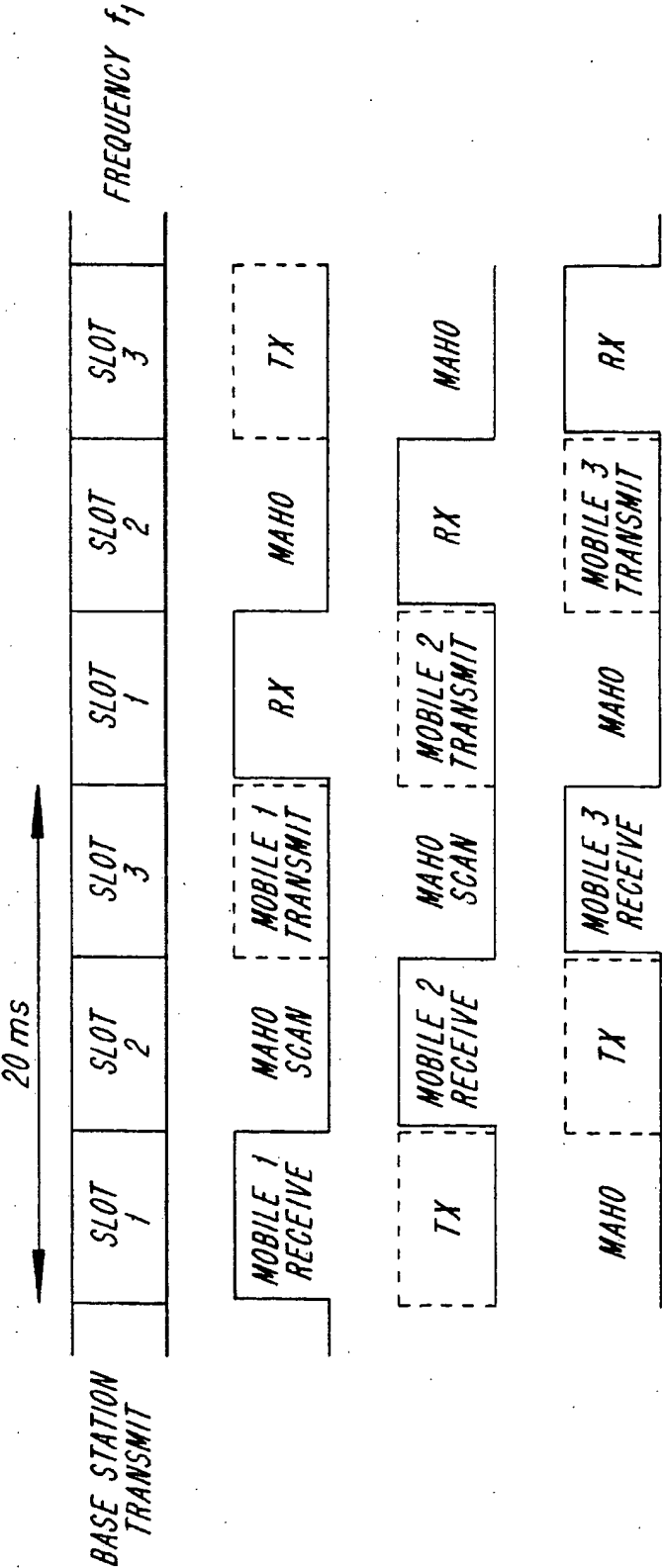


Fig. 2

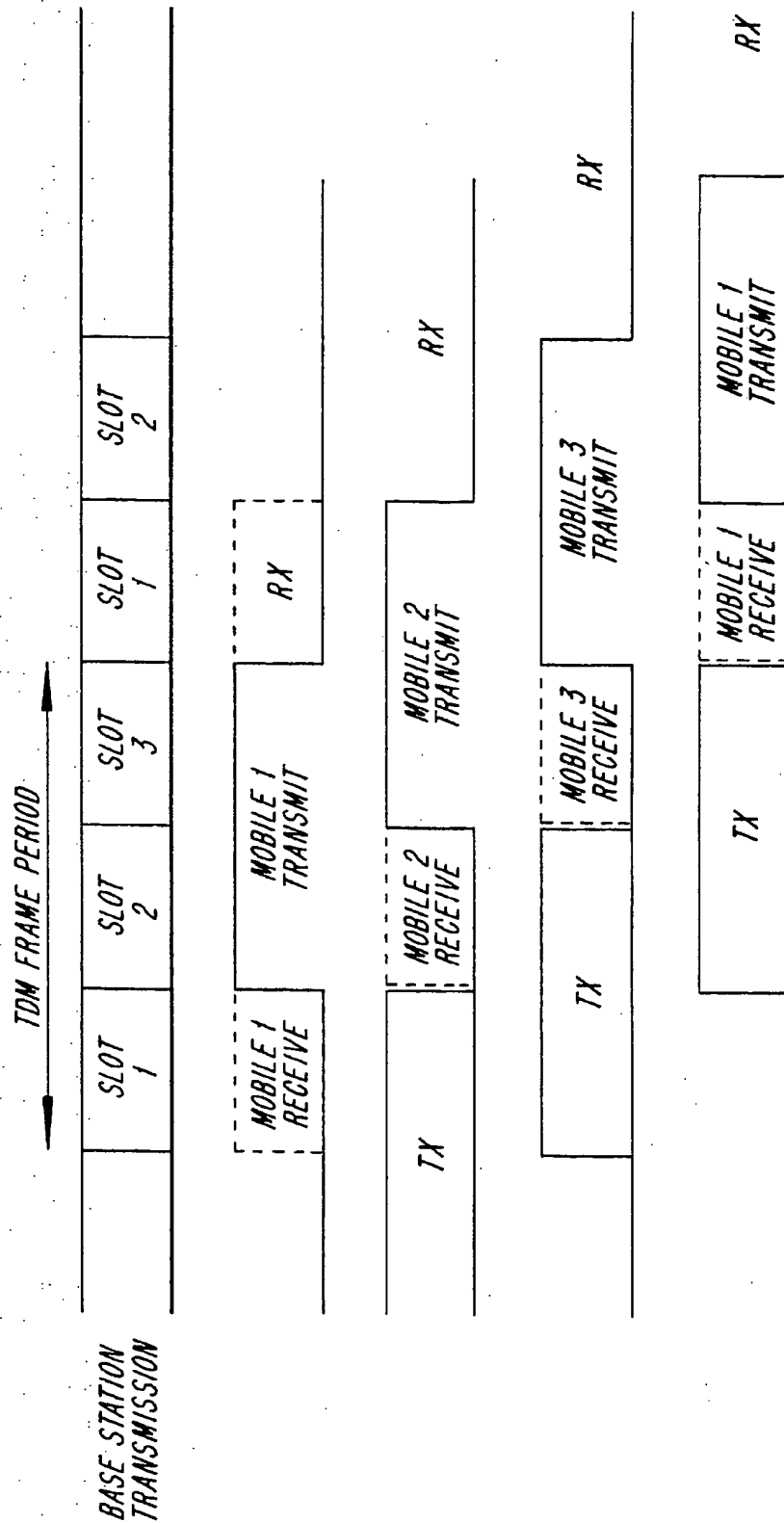


Fig. 3

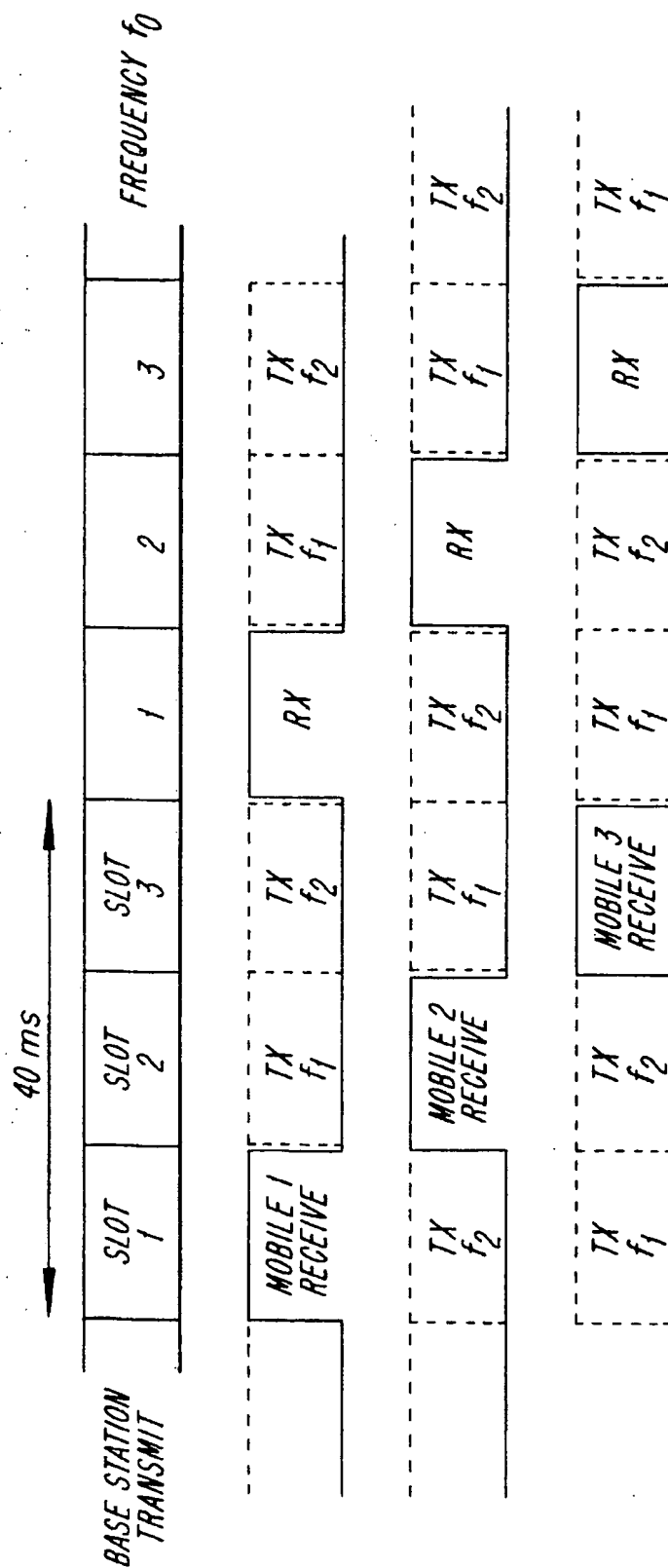


Fig. 4

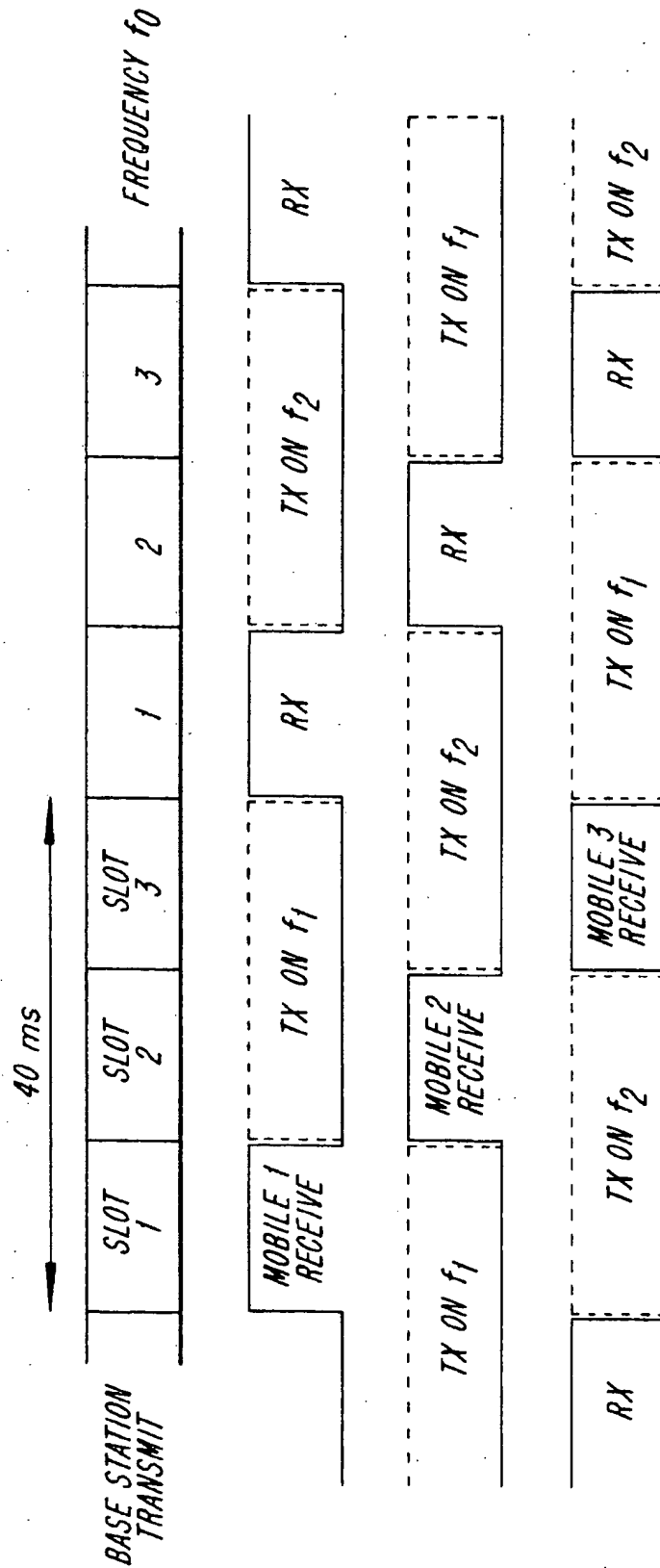


Fig. 5

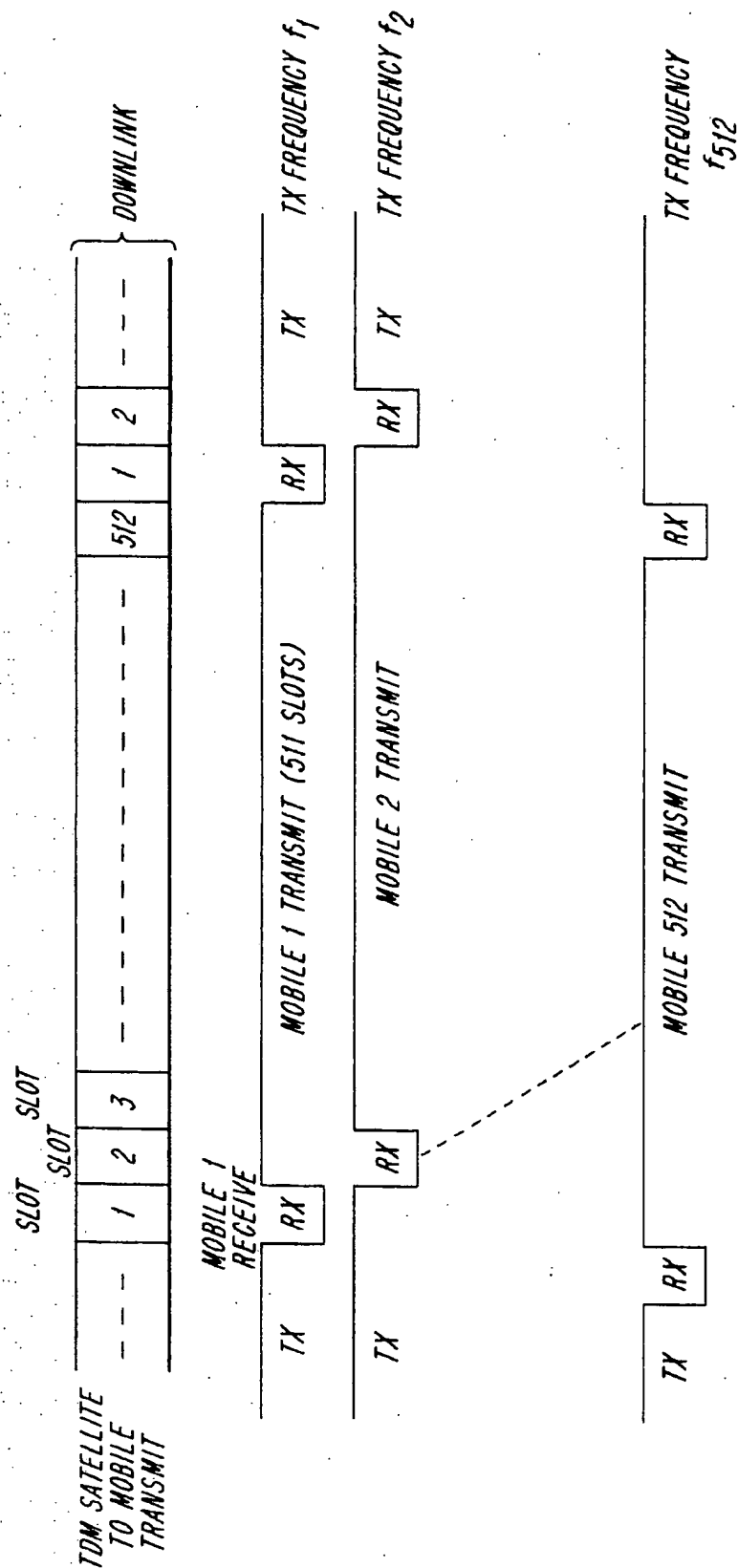


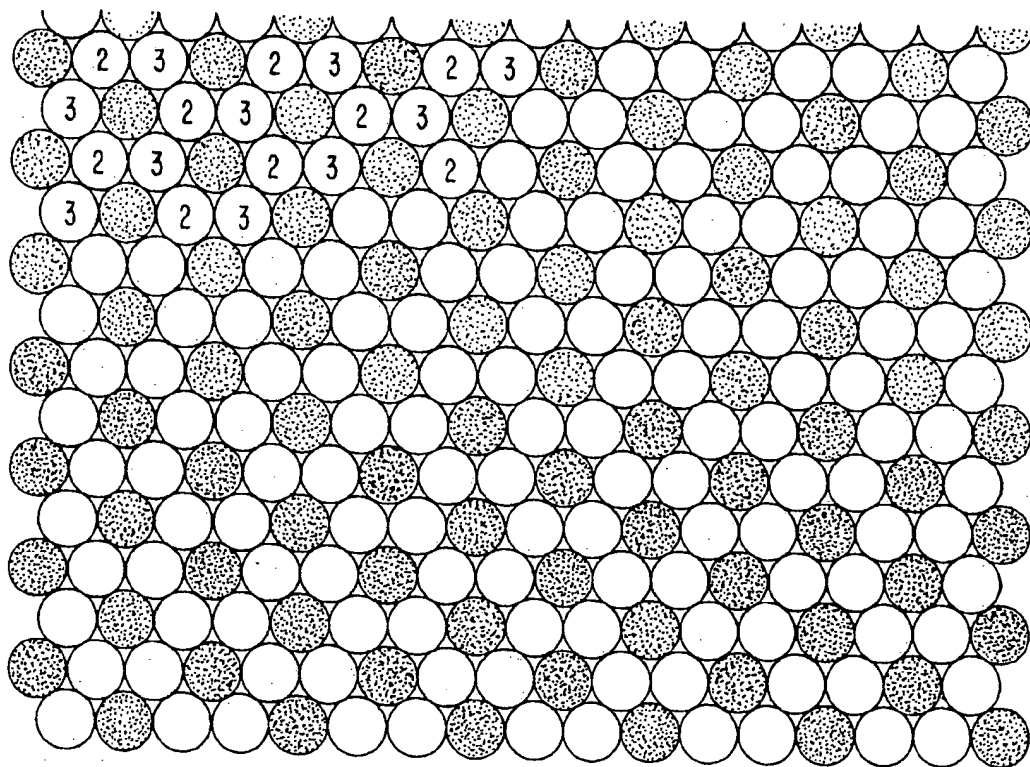
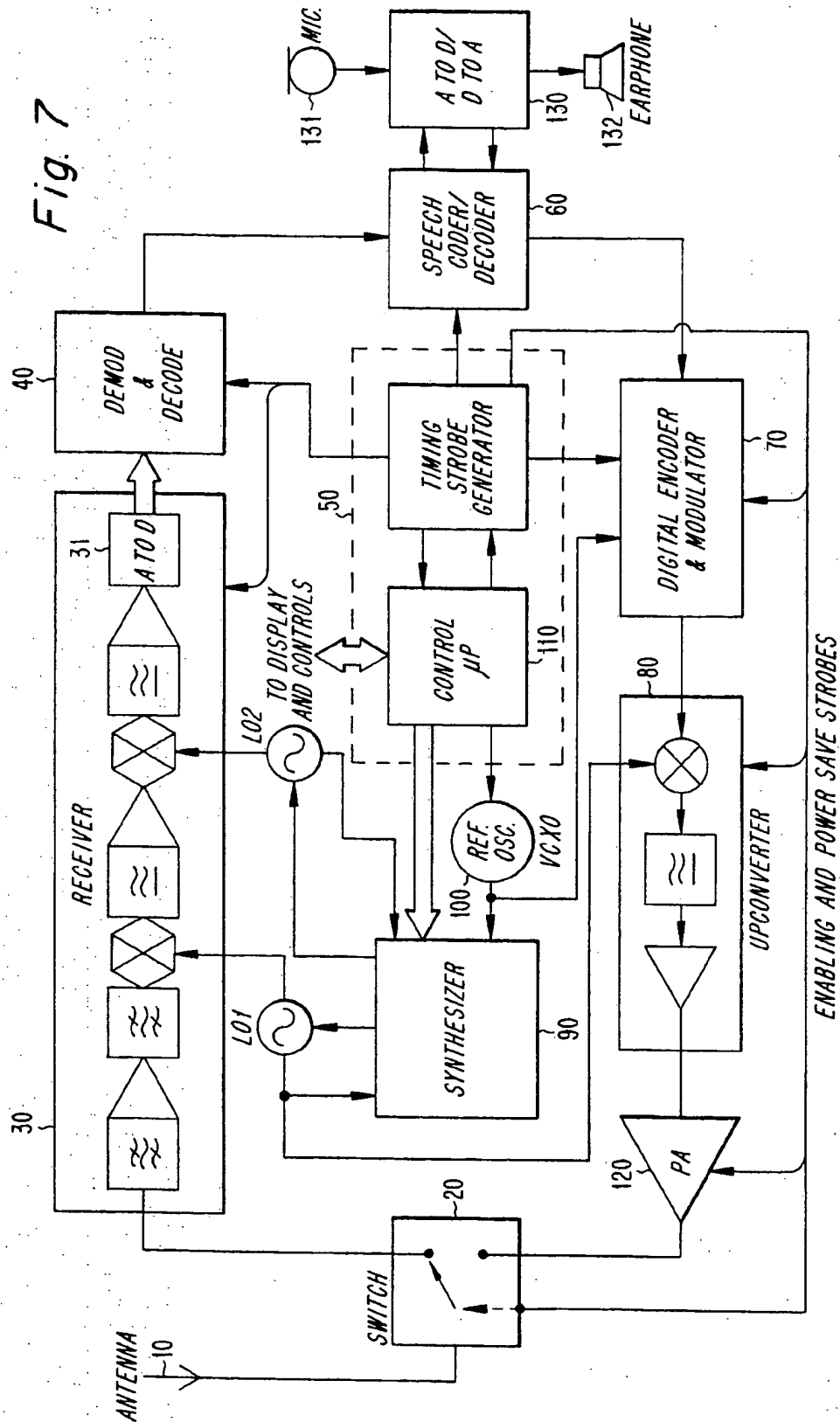
Fig. 6

Fig. 7



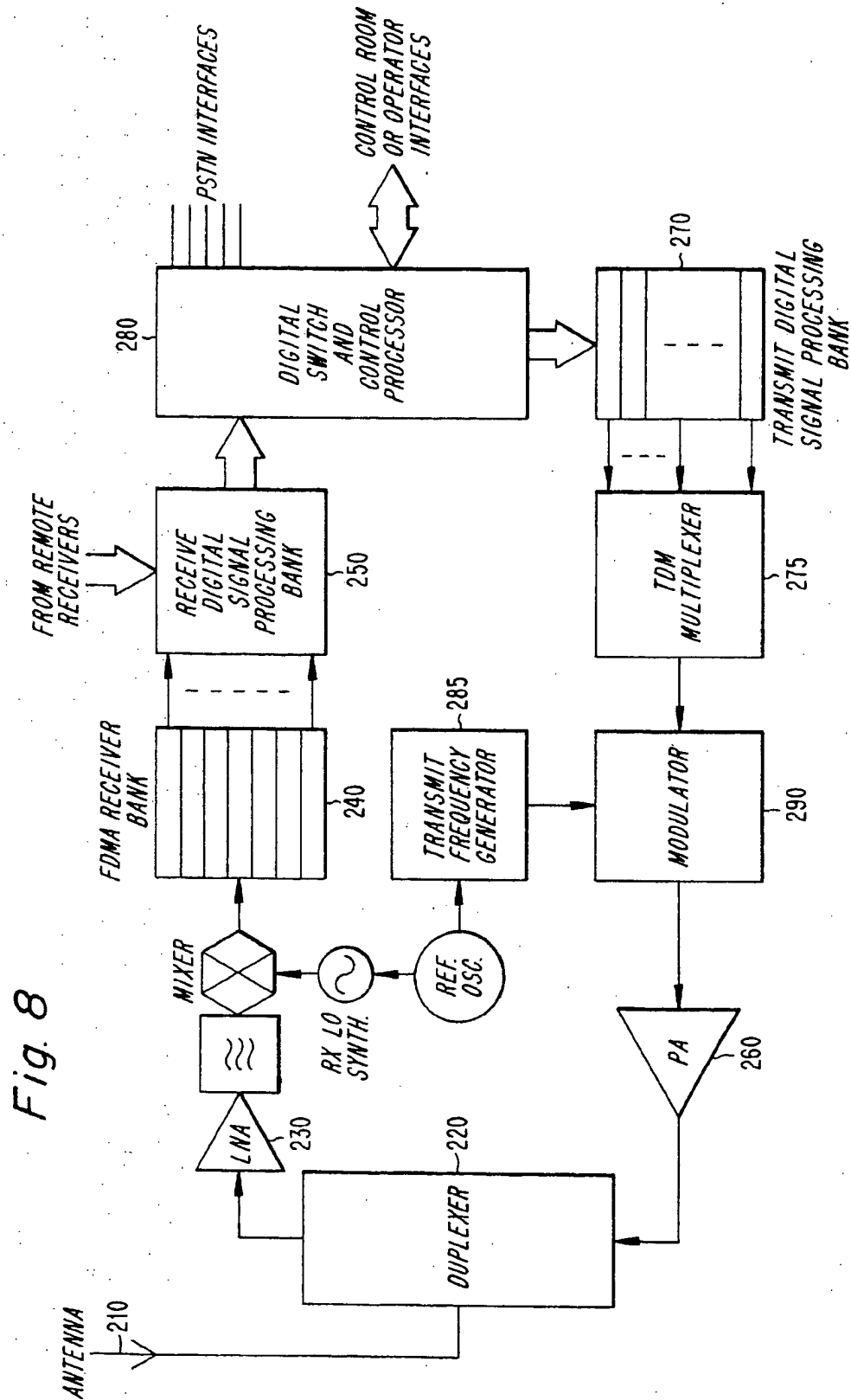
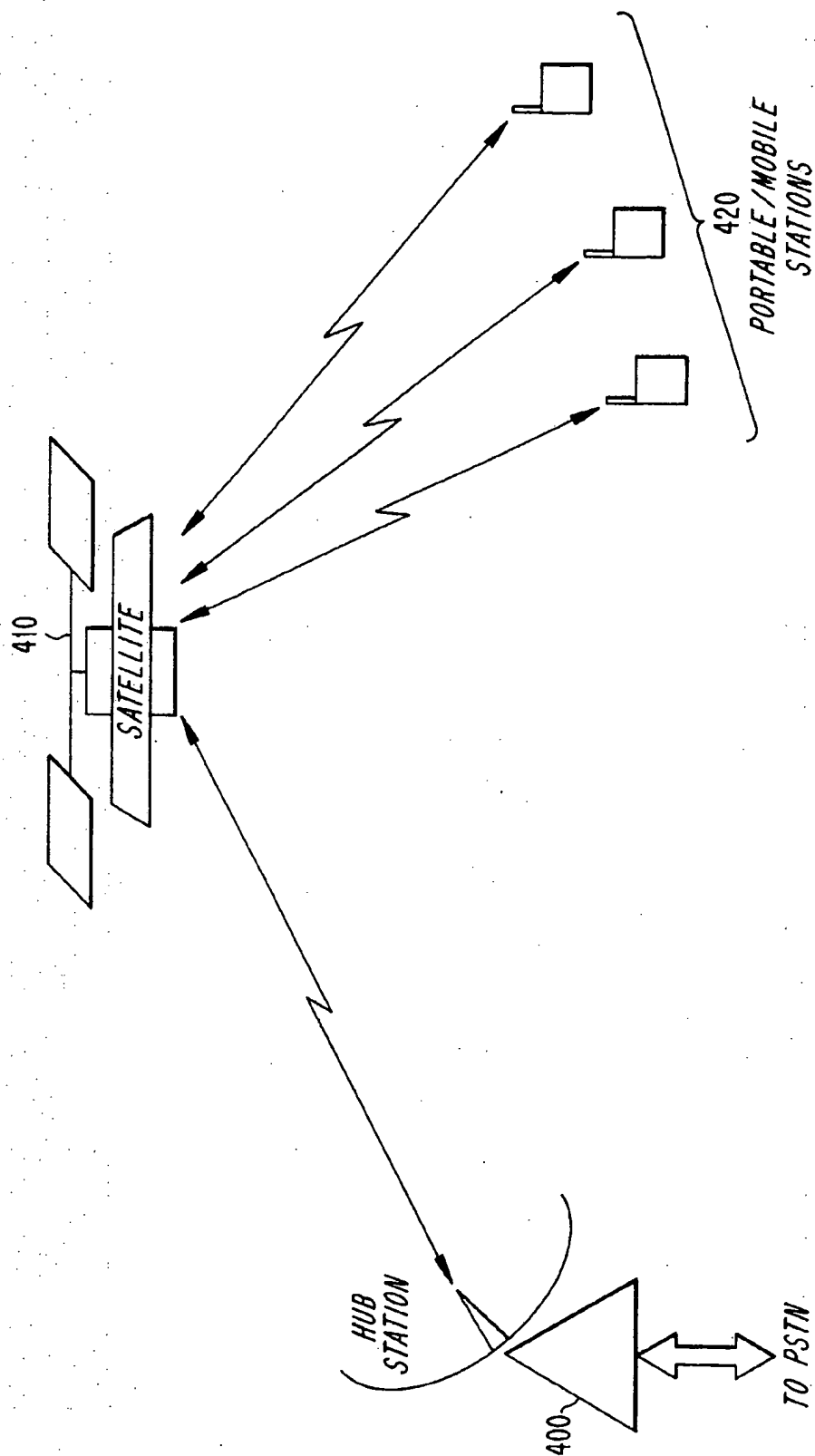


Fig. 9



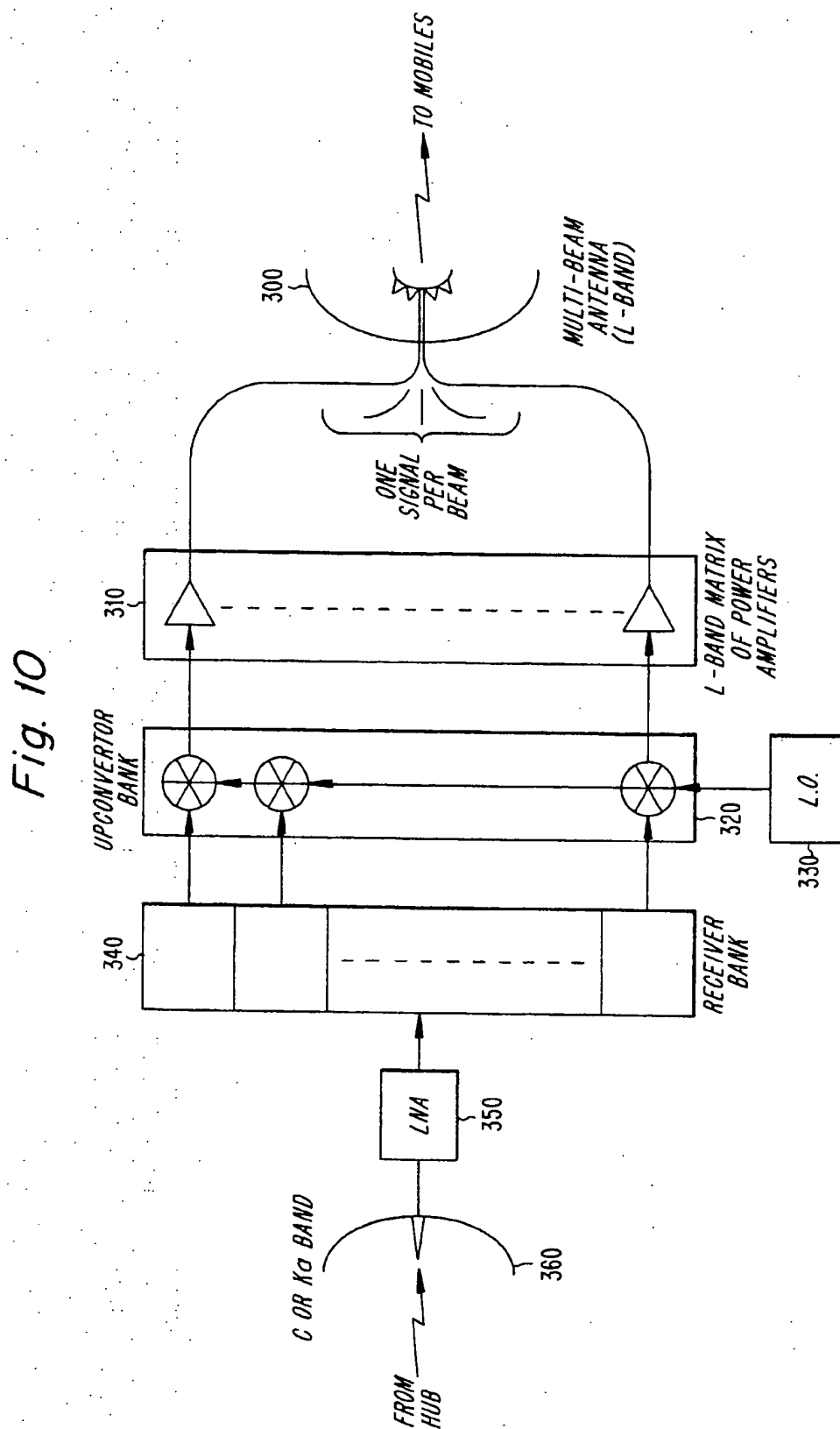
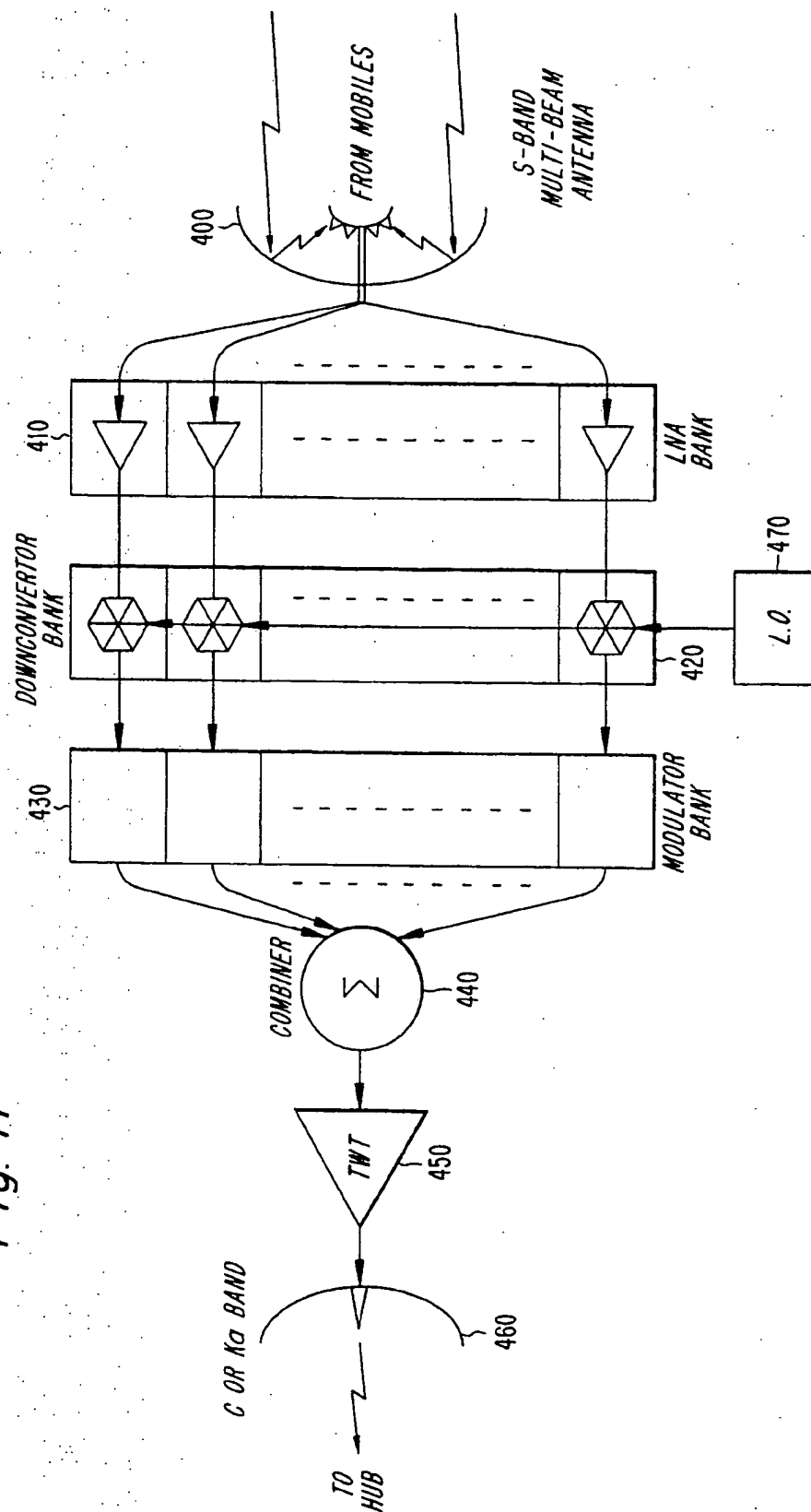


Fig. 11



TDMA/FDMA/CDMA HYBRID RADIO ACCESS METHODS

FIELD OF THE INVENTION

The present invention relates to duplex communication systems for cellular radiotelephone systems for simultaneous two way voice communications. The present invention is also relevant to certain simplex systems, for example, landmobile radio systems that provide "press-to-talk" communications facilities between a plurality of correspondents in a radio net. The present invention is also relevant to satellite communications with portable or mobile terminals.

BACKGROUND OF THE INVENTION

The prior art contains several examples of duplex radio communications using Frequency Division Multiple Access (FDMA) in which different radio telephones each have a unique pair of frequencies for transmissions in the transmit and receive directions, for example, the U.S. AMPS cellular telephone system. The prior art also discloses duplex radio communication systems using Time Division Multiple Access (TDMA) in which each radio telephone has a unique time slot on a first shared frequency for communication in one direction, and a second unique time slot on a second shared frequency for communication in the other direction, for example, the European GSM digital system or the U.S. Digital Cellular standard IS-54. In these systems, the time slots in the respective directions are furthermore offset in time from each other so that the portable radiotelephones do not need to transmit and receive simultaneously. This eliminates the need for a transmit/receive duplexing filter which is required by a radiotelephone operating in a FDMA system. Instead, a so-called "time duplex" phone, as envisioned in the prior European Cellular System GSM, uses a simpler transmit/receive switch to couple the antenna alternately to the receiver or the transmitter.

In certain applications, neither TDMA nor FDMA provides an optimal solution. The TDMA system requires higher peak transmitter power to compensate for compressing the transmission into a time slot that is only a fraction of the total time, since it is the mean power that governs the range and quality of the communication. This is not an issue for a base station that in any case must have enough transmitter power to support all mobile stations, and the total power is the same for FDMA and TDMA solutions. It is simpler and cheaper for the TDMA base stations to have one high power transmitter and one antenna which can be timeshared between all base/mobile links using Time Division Multiplexing (TDM). However, it is often inconvenient for TDMA mobile stations to generate high peak power. On the other hand, it is inconvenient for FDMA mobiles to use antenna duplexing filters. Therefore, the present invention seeks to provide a method of using TDM on the base-to-mobile link (downlink) combined with FDMA on the mobile-to-base link (uplink), while avoiding the need for a duplexer.

The prior art discloses examples of mixed TDM/FDMA systems, such as the British Army's PTARMIGAN Single Channel Radio Access System (SCRA). The SCRA system is in fact a military radiotelephone system, and uses TDM on the downlink on a first frequency band while using FDMA on the uplink by allocating a separate frequency in a second frequency band to each mobile uplink. The SCRA system, however, requires either separate antennas for the uplink and

downlink, respectively, or a duplexing filter to permit simultaneous transmission and reception through one antenna.

FIG. 1 illustrate the prior art transmission format described in the U.S. Digital Cellular standard IS-54. A base station transmits information continuously in frames of data which are 20 ms long. The data in question is composed of digitized speech information generated by a digital speech compression algorithm interspersed with synchronization, signalling and control symbols. Each 20 ms frame of data is divided into three time slots and each time slot contains information destined for one of three mobile stations. Thus, a particular mobile station only needs to turn on its receiver for one-third of the time since the data for the particular mobile station is confined to one of three time slots that make up the frame. In the reverse direction, the 20 ms frame is likewise divided into three time slots. Each mobile transmitter uses only one of the two time slots in which it is not receiving, which leaves the other third of the time which can be utilized to scan other base station frequencies to see if another base station is received more strongly. These signal strength measurements are reported over the uplink channel to the current base station, which makes a decision on whether to hand off communications with that mobile station to a stronger base station. Utilizing signal strength measurements performed by mobile stations in making handoff decisions is called "Mobile Assisted Handover" (MAHO).

In this prior art system, it can be seen that a mobile transmits for only one-third of the available time and therefore has to use three times the peak power that otherwise might have been sufficient if continuous transmission had been employed. If continuous transmission had been employed, all three mobile transmissions would be overlapping in time and would therefore have to be given different frequency channels as in the British Army's PTARMIGAN SCRA system. Furthermore, transmit/receive duplexing filters would be needed to allow simultaneous transmission and reception in the mobile station.

SUMMARY OF THE DISCLOSURE

The present invention relates to a radio access method for facilitating communication between at least one first station and a plurality of second stations. First, each signal intended for transmission is buffered at a first station. The signals are then divided into equal length segments. The signal segments intended for a particular one of the second stations is transmitted using a corresponding time slot in a regularly repeating time multiplex frame. The signal segments transmitted by the first station are received at at least one of the second stations and the signal segments are assembled from successive corresponding time slots to reconstruct said intended signal. A transmit frequency channel uniquely associated with the corresponding receive time slot is determined at the second station. Finally, a signal intended for transmission to the first station is buffered in the second station and compressed for transmission using the transmit frequency channel during substantially the entire time period that the second station is not receiving.

The present invention also discloses a radio transmitter/receiver for communicating in both directions with a radio network. The radio transmitter/receiver comprises a timing control unit for sequencing transmit and receive functions. An antenna switch alternately connects an antenna to the receiver and transmitter under the control of the timing control unit. A receiver portion controlled by the timing control unit receives a signal from the radio network during

an allocated time slot in the time division multiplex frame period. Finally, a transmitter portion capable of being controlled by the timing control unit transmits during the rest of the time division multiplex frame period when the receiver portion is not receiving.

The present invention also discloses a method of communication for providing telephone communications between at least one outstation and a subscriber in a public switched telephone network comprising at least one orbiting satellite and at least one ground control station in communication with the satellite and the public switched telephone network. First, a signal from the satellite bearing time division multiplex information comprised of time slots in a repetitive TDMA frame period is transmitted, wherein each time slot is allocated for reception by one of the outstations. After receiving signal bursts from the satellite, the outstations determine a transmit frequency channel uniquely associated with said allocated time slot. Finally, the outstation transmits signal bursts using the transmit frequency channel to the satellite using substantially all of the remaining TDM frame during which the outstation is not receiving.

BRIEF DESCRIPTION OF THE DRAWINGS

The present invention will now be described in more detail with reference to preferred embodiments of the invention, given only by way of example, and illustrated in the accompanying drawings in which:

FIG. 1 illustrates a prior art TDMA format;

FIG. 2 illustrates a TDM/FDMA format with overlapping mobile transmissions according to one embodiment of the present invention;

FIG. 3 illustrates a TDM/FDMA hybrid format according to one embodiment of the present invention;

FIG. 4 illustrates a TDM/FDMA hybrid format according to one embodiment of the present invention;

FIG. 5 illustrates the application of the present invention to satellite communications with a large number of time slots;

FIG. 6 illustrates a 3-cell frequency re-use plan;

FIG. 7 illustrates a block diagram of a portable radio according to one embodiment of the present invention;

FIG. 8 illustrates a base station for one embodiment of the present invention;

FIG. 9 illustrates satellite/mobile communications in one embodiment of the present invention;

FIG. 10 illustrates a hub-to-mobile satellite transponder; and

FIG. 11 illustrates a mobile-to-hub satellite transponder.

DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENTS

In a 3-slot TDMA communication system where the Mobile Assisted Handover feature is not required, but where simultaneous transmitting and receiving is to be avoided so as to eliminate the transmit/receive duplex filter, the present invention extends the transmit duty factor from one-third to two-thirds, thus halving the peak power requirement. The uplink and downlink formats according to the present invention are illustrated in FIG. 2.

As illustrated in FIG. 2, two of the three mobile transmissions overlap at any one time. In order to permit them to overlap in time, they must be made orthogonal, i.e., non-interfering, in some other domain such as the frequency

domain. Since using twice the time for transmission allows the transmission data rate to be halved, it is possible to accommodate two transmissions within the same bandwidth by arranging that one of the transmissions uses the top half of an allocated bandwidth while the other transmission uses the bottom half, or vice versa. For example, a first mobile station may use the upper half of the channel bandwidth and halfway through its two-thirds transmit period a second mobile station can start transmitting in the lower half of the channel. Then, after a further one-third of the frame period, the first mobile station will finish using the upper half of the channel and a third mobile station can begin transmitting in the upper half channel. After a further one-third period, the second mobile station will finish using the lower half channel, and the first mobile station can start transmitting again. However, the first mobile station will be operating on the lower half channel instead of the upper half channel it originally operated on. This problem arises only when an odd number of time slot are used in combination with channel bandwidth division by an even number, and may be solved by either of the two methods described below.

The solution in which two mobiles use the upper and lower half channels respectively for two-thirds of the time while a third mobile station uses the upper half channel for its first third of the time and then switches to the lower half channel for its second-third of the time is avoided since the aim of any solution should be that all mobiles function identically independent of time slot. It is preferable, if frequency switching midway through the transmit burst is to take place in any mobile station that it takes place in all mobiles so that the system has a uniform design.

FIG. 3 illustrates one embodiment of the present invention in which a first mobile station receives a first third of the base station transmission and then transmits to the base station using the upper half of the uplink channel for the first one-third of its two-thirds transmit period. After the first third transmit period, the first mobile station switches frequency so as to use the lower half channel for its second one-third transmission period. Meanwhile, a second mobile station has received the second third of the base station's 40 ms frame and starts transmitting in the upper half channel when the first mobile switches to the lower half channel. Then, when the second mobile station switches to the lower half channel midway through its transmit burst, a third mobile station begins transmission in the upper half channel. When the third mobile station switches to the lower half channel, the first mobile station begins transmitting again in the upper half channel. The mid-burst frequency switch from the upper to lower half channel is preferably accomplished not by means of a fast switching frequency synthesizer, but rather by means of applying a systematic phase rotation to the transmitted signal in order to provide a plus or minus frequency offset from the center of the channel. This can be performed within the digital signal processing used for generating the modulation waveforms as will be discussed below.

A second embodiment of the present invention, which avoids a mid-burst frequency shift, is illustrated in FIG. 4. Here, a first mobile station first transmits in the upper half channel and halfway through its two-third transmit frame a second mobile station starts transmitting using the lower half channel. Midway through the second mobile station's transmit period, the first mobile station finishes transmission and a third mobile station begins transmitting using the upper half channel. Midway through the third mobile station's transmit period, the second mobile station finishes its transmission on the lower half channel. At this time, the first

mobile station begins transmitting again on the lower half channel which is opposite from the channel it initially used. In this embodiment, every mobile functions identically but alternates between transmitting in the upper and lower half channels on successive bursts. In this system, the one-third receive period of 13.3 ms between successive bursts is available for changing the transmit frequency, which can be accomplished by using a frequency synthesizer with a modest frequency changing speed.

It will be appreciated that the invention is not limited to a system with three time slots. When an even number of time slots is used, for example four, and the mobile transmit time is three-fourths of a frame, the transmissions of three mobile stations overlap in frequency at one time. In such a situation, the channel bandwidth can be divided into three, and each mobile station can then use the three subbands in sequence. Alternatively, one mobile can transmit for $\frac{3}{4}$ of the time in $\frac{1}{2}$ of the bandwidth, while the other three use the other combinations of half-frame period and half-channel-bandwidth.

The above described solutions are in general characterized by dividing the uplink channel bandwidth into a number of subbands which is at least one less than the number of downlink timeslots 'N'. For example, the three-timeslot case divides the channel into upper and lower halves, while the four-timeslot case divides the channel into three subbands. This is compatible with the reduction in bit rate by the factor (N-1) when the transmitter operates for N-1 timeslots instead of just one timeslot. When N is small, it is difficult to divide the number into N sub-bands without also reducing the bit rate by N instead of only N-1. For example, in the three-timeslot system, it would be difficult to accommodate a half-bit rate transmission, obtained by transmitting only $\frac{1}{2}$ of the time instead of $\frac{1}{3}$, in only $\frac{1}{3}$ of the bandwidth. This difficulty however disappears when N is large.

FIG. 5 illustrates one embodiment of the present invention which may be advantageous for satellite-mobile communications. In this embodiment, a 512-slot TDM downlink is combined with a 512-subband FDMA uplink. To avoid using a duplex filter in the mobile station, the signal for transmission is compressed into the $\frac{1}{512}$ of the time remaining over after the mobile has received its $\frac{1}{512}$ of the downlink TDM format. However, the 0.2% increase in the information rate does not hinder it from being accommodated in $\frac{1}{512}$ of the bandwidth. Without this signalling format, either the mobile would have to transmit and receive at the same time, necessitating a duplex filter which results in undesirable signal losses, or else use TDMA on the uplink involving the mobile transmitting, for example, for $\frac{1}{512}$ of the time using 512 times the peak power which results in an undesirable increase in the peak power and current requirements from the power supply. The present invention may, of course, also in such a case permit transmission for $\frac{1}{512}$ of the time or even less, without severe difficulties, the compression of the information not being restricted to clearing one timeslot for receiving if there are other demands on time, such as providing guard times between transmit and receive.

It can also be advantageous to consider other hybrids of TDMA and FDMA and even CDMA for the uplink in the case of satellite communications. Satellites orbiting at altitudes which are lower than the geostationary altitude exhibit significant velocities relative to stationary or mobile terminals on the ground. This can result in the Doppler shifts of the frequency received at the satellite from the ground terminals that are significant compared with the narrow

transmission bandwidths of a pure FDMA uplink. It is therefore sometimes desirable to increase the uplink bandwidth to make Doppler shift relatively insignificant, without reducing the capacity of the system. A small factor increase, such as 2, 4 or 8 may often suffice. One method of accomplishing a 2:1 increase in bandwidth while accommodating the same number of transmitters would be for each uplink transmission to be compressed into $\frac{1}{256}$ of the time on one of 256 available subbands; two transmissions in each subband would then be accommodated by TDMA through a first mobile station's transmitter using the $\frac{1}{512}$ timeslots numbered 1 to 256 and a second mobile station's transmitter using timeslots 257 to 512. The first mobile station receives on timeslot 257, while the second mobile station receives, for example, on timeslot 1, thus avoiding the need for being able to simultaneously transmit and receive. This principle may be extended to 4 timeslots on each of 128 subbands, or 8 timeslots on each of 64 subbands, and so on. However, The mobile station's peak transmitter power must be increased as its duty factor is reduced by using more TDMA and less FDMA on the uplink.

The bandwidth may instead be expanded by the use of Code Divisional Multiple Access (CDMA) on the uplink. In CDMA, each of the original information bits is transmitted a number of times with or without a polarity inversion according to the bits of an access code. For example, a four-times increase in bandwidth is achieved by using the access code 1100 and transmitting, in place of the original bit B1, the sequence B1B1B1B1; B2 is also replaced by B2B2B2B2 and so on, giving a fourfold increase in bit rate. Another mobile transmission may be allowed to overlap this by use of a different access code, preferably an orthogonal code such as 1001. The other mutually orthogonal codes are 1111 and 1010, resulting in four, overlapping, non-interfering transmissions sharing a four times wider subband. This achieves a fourfold increase in uplink signal bandwidth which is desired to render Doppler shift relatively insignificant while preserving capacity without requiring higher peak transmitter powers from the mobile station.

The capacity in a cellular telephone system or in a mobile-satellite communications system, depends on being able to re-use the limited number of allocated frequencies for more than one conversation. The service area to be covered is usually divided into a number of cells each served by a base station (or illuminated by a satellite antenna spot-beam). Ideally, it should be possible to utilize the whole of the allocated spectrum immediately in each adjacent cell, however this is not conventionally possible due to the interference of neighboring cells used in the same frequencies. As a result, a frequency re-use plan must be employed to control interference levels. For example, a so-called 3-cell frequency re-use plan may be employed, as illustrated in FIG. 6. A 3-cell frequency re-use plan guarantees a certain minimum desired Signal-to-Interference (C/I) ratio that may suffice if adequate error-correction coding is employed on the transmitted signal. In general, the C/I is better for a 3-cell re-use plan in the satellite case than in the ground cellular case, due to the sidelobes of the satellite's cell illumination profile tapering off more rapidly out-of-cell than the signal strength reduction with increasing distance in ground propagation.

A problem can arise in applying frequency re-use plans to a TDM downlink. A limited allocated frequency spectrum has to be divided into three to permit a 3-cell re-use plan. As a result, the bandwidth of a full TDM solution can no longer be accommodated. This problem is solved according to an aspect of the present invention, by using a time-reuse plan

instead of a frequency reuse plan on the TDM downlink, combined with a corresponding frequency reuse plan on the FDMA uplink.

In the time-reuse plan, cells designated '1' in FIG. 6 are illuminated from the satellite or from their respective ground base stations using the first $\frac{1}{3}$ rd of the timeslots in the TDM format, using the full available frequency spectrum. The cells labelled '2' then receive illumination during the second $\frac{1}{3}$ rd of the TDM format, and so-on. In this way, adjacent cells are not illuminated with the same frequencies at the same time, but the full TDM signal bandwidth is still transmitted. For example, in a 512 timeslot TDM format, cells numbered '1' are illuminated for the first 170 time slots. Each mobile terminal, after receiving its respective $\frac{1}{3}$ 12th time slot may transmit for the remaining $\frac{2}{3}$ 12ths of the frame using a designated one of the first 170 out of 512 uplink FDMA channels. The cells numbered '2' are then illuminated for the second 170 out of the 512 timeslots, and corresponding mobiles in those cells reply using FDMA and respective uplink channel frequencies 171 to 340. The cells numbered '3' then become illuminated for the third 170 out of 512 timeslots, and their mobiles reply on uplink frequencies 341 to 510. The remaining two timeslots can be reserved for illuminating all cells with a special signal used for paging and call set-up. Likewise, the two correspondingly unused uplink channel frequencies can be reserved for mobiles wishing to initiate contact with the system by performing a so-called random access.

By using the above system of time re-use plans on the downlink combined with a matching frequency reuse plan on the uplink, the described TDM/FDMA hybrid access method can be employed while controlling the interference levels between adjacent cells.

As previously disclosed, it can be desirable to widen the uplink channel bandwidths by employing TDMA or CDMA on the otherwise FDMA uplink and reducing the number of FDMA channels commensurately, in order to correct for Doppler shifts. It will be appreciated by those of ordinary skill in the art that the actual numbers used above are exemplary and do not imply a restriction of the present invention to those examples.

FIG. 7 illustrated a preferred implementation of a mobile or portable radio suitable for use in the present invention. An antenna 10 which operates at both the uplink and downlink frequencies is connected alternately to receiver 30 and transmit power amplifier 120 by means of a T/R switch 20, which is controlled by the TDM timing generator 50. In the alternative, transmit/receive duplexing filters can be used if uplink and downlink frequencies are sufficiently separated to permit simple filters with low loss. When uplink and downlink frequencies are widely separated, a single antenna may not be efficient, in which case separate transmit and receive antennas may be necessary. This does not, however, change the principle of the present invention, which is to avoid having the transmitter active during receiving instances.

The timing generator supplies timing and control pulses to the switch 20, the receiver 30 and the digital demodulator and decoder 40 in order to provide them with power to select the signal in allocated timeslots on the downlink. The receiver 30 has sufficient bandwidth to receive the entire TDM downlink signal spectrum, but only one timeslot per 40 ms frame of this bit rate stream is selected for processing in the digital demodulator and decoder 40. During this selected timeslot, the signal from the receiver is digitized in A to D converter 31 and recorded in a buffer memory contained in the demodulator 40. The digitization technique

preferably preserves the complex vector nature of the signal, for example, by splitting the real(I) and imaginary (Q) parts by means of a quadrature mixer, and then digitizing each part. An alternative to this so-called I, Q, or Cartesian method is the LOGPOLAR method described in U.S. Pat. No. 5,048,059 which is assigned to the same assignee and is hereby incorporated by reference. Another alternative technique is the co-called homodyne or zero-IF receiver such as described in U.S. patent application Ser. No. 07/578,251 now U.S. Pat. No. 5,241,702 which is hereby incorporated by reference.

The complex vectors recorded in the buffer memory are then processed by the digital demodulator and decoder 40 during the rest of the frame time before collection of the next timeslot's complex signal samples. The demodulation stage of the processing can involve channel equalization or echo-cancelling to mitigate the effects of multipath propagation. Typical algorithms suitable for this are disclosed in U.S. patent application Ser. No. 07/964,848 and U.S. Pat. No. 5,331,666 assigned to the same assignee which are incorporated herein by reference.

To help bridge fading, error-correction encoded data frames may be spread over more than one timeslot by means of interleaving, so that a number of timeslots have to be collected and deinterleaved before the first frame of speech data is error-correction decoded. The demodulation of the signal samples in each timeslot should preferably be optimized together with the error-correction decoding algorithm for best performance at low signal-to-noise ratios, for example, by passing soft-decision information from the demodulator to the decoder, or by combining demodulation with decoding in a so-called decodulator.

After demodulation and error correction decoding, using for example, a soft-decision based convolutional decoder, a frame's worth of error-correction decoded speech data is passed to the speech coder/decoder 60, where it is turned into PCM speech samples at 8 kilo samples/second using a decoder that matches the encoder at the originating transmitter. The speech coding/decoding technique can for example be Residual Pulse Excited Linear Predictive Coding (RPLP) or Code Book Excited Linear Predictive Coding (CELP) which compresses an 8 kilo sample/second PCM voice signal down to 4.2 kilobits/second at the transmitter, and conversely expands the 4.2 kilobits/second signal from the decoder 40 to 8 kilobits/second PCM signal again for D to A conversion in D to A converter 130 and audio amplification for driving an earphone 132.

In principle, the receiver only needs to receive a single frequency on which all signals from all mobiles are multiplexed and modulated. As a result, the receiver does not have to tune to alternative frequencies, but rather, selects between all available timeslots. The control microprocessor 110 receives information on a calling/paging slot during call set-up designating the slot to be used for the call. The control microprocessor 110 then programs the timing generator accordingly, to generate all control pulses necessary to power the receiver and transmitter on and off according to the inventive TDM/FDMA hybrid formats described herein. The control microprocessor also programs the transmit synthesizer 90 to generate an FDMA channel uplink frequency associated with the allocated downlink timeslot with the help of upconverter 80. The upconverter 80 can operate in several manners. First, the upconverter 80 can operate by mixing a fixed modulated frequency (TX IF) with the variable frequency produced by a programmable frequency synthesizer to generate a sum or difference frequency at the desired transmit channel frequency, wherein either the sum

or the difference is selected by a filter. In the alternative, the upconverter 80 can operate by mixing a signal from a voltage controlled oscillator with the synthesizer frequency to produce a difference frequency that is phase-compared to the fixed modulated frequency in a phase error detector, the phase error then being amplified and applied to the VCO in order to lock it to the modulated TX IF, thus causing the VCO phase to follow phase modulation on the TX IF. The determination of which method to select depends on whether the selected transmit modulation technique is pure phase modulation, i.e., constant amplitude modulation, or whether the selected modulation involves a varying amplitude component.

In the reverse direction, a speech signal from the microphone 131 is first amplified and converted to 8 kilobit sample/second PCM using A to D converter 13 and then compressed to a reduced bit rate using the speech coder 60. Speech compression techniques such as RELP and CELP that compress speech to as low as 4.2 kilobit samples/second generally operate on 40 ms frames of speech samples at a time. A frame is typically compressed to 163 bits, that are then error-correction coded in digital encoder 70 before being modulated onto a radio frequency. The modulated radio frequency may be a fixed intermediate frequency locked to an accurate reference oscillator 100. The signal is then upconverted in upconverter 80 by mixing with transmit synthesizer 90 to the final uplink frequency signal which is then amplified by transmit power amplifier 120 and passed by the switch 20 to the antenna 10. The digital encoder and modulator 70 includes buffering (and if used, interleaving) in order to compress the transmission into the available time left over after receiving the downlink timeslot, so as to implement the aspect of the present invention, thus avoiding simultaneous transmitting and receiving. The timing of this compression, modulation, and activation and deactivation of the power amplifier at appropriate instances is also controlled by timing generator 50 so as to achieve the coordination between transmit and receive timing.

In some applications, the receiver may have to be able to tune to alternative channel frequencies as well as to select between the TDM timeslots on those channels. In those instances, the receiver 30 would also contain a frequency synthesizer programmed by control microprocessor 110 and locked to the accuracy of the reference frequency oscillator 100. The allocated frequency is given over a calling or paging channel at call set-up.

In landmobile radio applications using a push-to-talk operation, the mobile terminal may be a member of a group or not of correspondents sharing a trunked radio system with other groups. In trunked systems, all idle radios listen to a call set-up channel. When a radio transmits by activation of the talk switch, a short message is transmitted on the corresponding call set-up uplink channel requesting a channel allocation. The receiving base station network immediately replies on the downlink call set-up channel with a currently idle frequency/timeslot allocation to which the mobile terminal then adapts for the rest of the transmission. When the press-to-talk switch of the transmitting radio is released, an end of message signal is transmitted to expedite the reversion of the base network and other members of the group to idle mode in which they listen to the call set-up channel. This procedure is fast and automatic, within a fraction of a second, so that it is completely unseen to the human operators.

In cellular or satellite telephone applications, idle mobile terminals listen to a particular timeslot/frequency designated by the calling/paging channel. Moreover, transmissions on

the calling/paging channel timeslot may be further submultiplexed to form less frequently repeated slots each associated with particular groups of mobiles, designated, for example, by the last few digits of their respective telephone numbers. These so called sleep mode groups are only paged in a particular submultiplex slot, which control microprocessor 110 is able to identify from received data and can thus program the timing generator 50 to wake up the receiver only at these instances, which results in considerable standby power current consumption savings.

Furthermore, the digital demodulator/decoder 40 can, after processing of each new received timeslot, produce an estimate of the frequency error of the receiver caused by inaccuracies of the reference oscillator 100 as well as Doppler shift, which can be significant in satellite systems. By using broadcast information from a satellite, the microprocessor 110 can correct for the Doppler shift and determine the error due solely to reference oscillator 100. The microprocessor 110 can then correct the error by sending a correction signal such as a tuning voltage to the oscillator, in order to insure that the transmit frequency which is referenced to the reference oscillator by transmit frequency synthesizer 90 is accurately generated. The process of correcting for Doppler shift involves determining the position or bearing relative to the satellite's orbit by utilizing any or all of the following parameters: measured rate-of-change of Doppler shift; satellite and antenna beam identification signals; broadcast information on the satellite's instantaneous three dimensional coordinates; previous mobile terminal position; elapsed time since last position estimate; and mobile terminal velocity.

In addition to the frequency correction mechanism mentioned above, the demodulator produces information on the position of signal samples in the buffer memory deemed to correspond to known sync symbols, which yields information on the accuracy of the timing produced by the timing generator 50. The microprocessor 110 performs sanity checks on this information and then uses it, if deemed valid, to demand small timing corrections of the timing generator 50 so as to correct for any drift.

FIG. 8 illustrates a block diagram of a base station implementation suitable for use in the present invention. A common antenna 210 is connected to a receiver low noise amplifier 230 and a transmit power amplifier 260 by a duplexing filter 220. The low noise amplifier passes the entire uplink frequency band to a bank of FDMA channel receivers 240. After digitizing the signal in each channel using one of the aforementioned complex vector digitizing techniques, the signals are processed in a bank of receive Digital Signal Processing devices 520 in order to perform demodulation and equalizing, error correction decoding, and speech decoding for each active channel. The resulting 8 kilobit samples/second speech signals are then time multiplexed using a standard digital telephone standard such as the T1 format for convenient connection to a digital switch or exchange 280 such as the ERICSSON AXE switch.

An alternative to the analog implementation of the FDMA receiver bank is to digitize the entire composite signal and to process it digitally to separate the individual FDMA signals. This is practical so long as the signal strength differences between the signals are not too great for the A to D converter's dynamic range. Another aspect of the present invention is the inclusion of power control means to restrict the signal level differences between different FDMA signals in order to facilitate digital implementation of the FDMA receiver filter bank with potential simplification of the base station. The power control means proposed is known to the

prior art and is based on the mobile stations assuming correlation between the signal strength they receive from the base station and the signal strength the base station receives from them. Thus, an increase in signal strength received from the base station is acted upon by a mobile station by reducing its transmitter power and vice versa. This is complemented by a slower power control means at the base station which includes up/down power control information in the signalling samples interleaved with voice symbols in each mobile station's timeslot.

The exchange/switch 280 selects between uplink signals received from the public switched telephone network or signals from an operator or control room for transmission on the downlink, according to call set-up information, requested routing, or preset information. The switch 280 supplies the selected signals multiplexed together according to some known digital telephone trunk format such as T1 and delivers the signals to the transmit DSP bank 270. The transmit DSP bank separately encodes each speech signal in the multiplex stream using a voice compression algorithm such as RELP or CELP. The transmit DSP bank then error correction codes the signals and remultiplexes the signals into the downlink TDM format for modulation using modulator 290 onto the downlink radio frequency and amplification using high power amplifier 260. The switch 280 also extracts call setup information for the uplink calling channels and inserts call setup information corresponding to paging slots in the downlink TDM format. This information is identified as data and not as speech to the respective DSP devices so that it bypasses RELP coding and instead is subjected to a more powerful form of error correction coding.

A land based system may furthermore include separate antennas and associated receive signal processing in order to effect space-diversity reception to improve range and to combat fading. The combination of signals processed from remote antennas with signals processed with antenna 210 can take place either within the demodulation and equalizing algorithm, or by means of a simple diversity selection on a speech frame by speech frame basis according to signal quality. Likewise, in the transmit direction, a second distant transmitter may receive signals from transmit DSP bank 270 for transmitting on the same frequency so as to improve area coverage. The mobile receiver illustrated in FIG. 7 is able by means of its equalizing demodulator algorithm to perceive delayed signals received from a second transmitter as echoes of the first transmitter and to utilize these signals in order to improve reception.

FIG. 9 illustrates a block diagram of satellite communications system for one embodiment of the present invention. An orbiting satellite 410 is in communication with at least one ground station or outstations called the HUB 400 as well as with a number of portable mobile phones 420. The phones are each serviced by an appropriate antenna beam from a multiple spot-beam antenna on the satellite providing high gain in the direction of each phone. The HUB communicates with the satellite using, for example, C-band or Ka-band frequencies, while the satellite communicates with the phones using, for example, L-band (uplink) and S-band (downlink) frequencies. In most cases, most calls will be between satellite phones and ordinary phones belonging to the public switched telephone network. The HUB station accepts calls from the PSTN and relays them to the mobile phone via the satellite, and conversely accepts calls from the mobile phones relayed from the satellite and connects them to the PSTN. A small percentage of calls can be mobile to mobile calls, and the HUB directly connects them to each

other without necessarily involving the PSTN. In some systems, two or more HUBS located in different parts of the world communicate with the same satellite. In this case, mobile to mobile calls may involve Hub-to-Hub connections which can be accomplished through international trunk lines that may be part of the PSTN system. Alternatively, the satellite HUB links can allocate some capacity for Hub-to-Hub communication via the satellite for such occurrences thus avoiding landline tariffs.

FIGS. 10 and 11 illustrate a satellite communications payload suitable for one embodiment of the present invention. FIG. 10 illustrates the downlink to the mobile phones while FIG. 11 illustrates the uplink from the mobile phones. Referring now to FIG. 10, an antenna 360 receives a number of signals from the HUB which are demodulated or coherently downconverted using a bank of receivers 340. The receiver output signals are then coherently upconverted in a bank of upconvertors 320 by mixing with a common local oscillator 330. The upconverted signals are now at the downlink frequency and are amplified by a bank of power amplifiers 310, wherein each amplifier is coupled to one element, a group of elements, or a feed of a multi-beam antenna or phased array. In one embodiment of the present invention, the amplifiers are class C transmit power amplifiers operated at maximum efficiency. In one embodiment of the present invention, the satellite transmitter comprises saturated travelling wave tubes. The HUB is thus able by sending appropriate signals to the satellite antenna 360 to determine what signals will be broadcast by a multi-beam antenna 300 at what time and in what direction. In this manner, it can be determined, for example, that in any particular time slot of the down TDM format only a subset of regions of the earth receive the signals, the regions being sufficiently separated in boresight angle so that they do not suffer interference from one region to another. In this way, independent signals can be sent to one phone in each region in each timeslot without interference. In the next timeslot, a different set of regions, i.e., those in between the first set of regions, are illuminated so that all regions receive the signal from some timeslots in the frame. Copending U.S. patent application Ser. No. 08/179,953, entitled "A Cellular/Satellite Communication System With Improved Frequency Re-use", filed Jan. 11, 1994, which is incorporated herein by reference, discloses how one to one re-use can be used for the present embodiment wherein every timeslot is used in all of a number of sub-regions.

When the system is operating at less than full capacity, not all of the timeslots in the frame will be active. Moreover, one half of a two party conversation is generally silent at any time so that an advantage can be gained by turning off the signal in the corresponding timeslot momentarily. When the number of timeslots is large, i.e., 512, it is statistically accurate to assume that only approximately 50% will be active at the same time. The power amplifiers 310 are arranged to draw little or no current during inactive or unallocated timeslots so that the mean consumption from the satellite prime power supply corresponds, even when fully loaded, to only half the power amplifier peak power consumption. For a given size solar array, the power amplifier peak power can thus be dimensioned to twice the value which the solar array otherwise would support.

Furthermore, peak capacity is reached only at certain times of the day, whereas the solar array converts the sun's energy into electrical power during a full 24 hour period. By using a rechargeable battery to average the power consumption in 24 hours, a further factor increase in peak transmitter power can be made relative to the continuous load the solar

array can support. An advantage of TDM downlink used in the present invention is that current consumption reduces in direct proportion to the under-utilization factor, in contrast with an FDMA or CDMA downlink which use power amplifiers which only reduce their current consumption by the square root of the under-utilization factor, if at all. Therefore, using a TDM downlink allows the full benefit to be taken of the average under-utilization factor.

In one embodiment of the present invention, the active time slots of any TDM signal are packed together to occupy adjacent time slots in a subframe period which is a portion of the TDM frame period. The inactive time slots form the rest of the TDM frame period. The subframe of any TDM signal retransmitted in one of the multiple satellite antenna beams does not overlap the subframes of the TDM signals transmitted in the neighboring beams.

Referring now to FIG. 11, a multi-beam antenna or multi-element phased array 400 receives signals on the uplink frequency from a plurality of mobiles. Mobiles in the same region of the earth use different FDMA channel frequencies on the uplink and according to the invention do not transmit during their received timeslots on the TDM downlink. Mobiles in a different region of the earth use the same set of frequencies as mobiles in the first region, therefore the antenna 400 receives a plurality of signals on each FDMA channel that arrive from different directions. In the case of a multi-beam antenna such as a parabola with space feeds, the different directions correspond to different beams so that signals on the same frequency appear in different beams and can thus be separated. This may require that adjacent beams do not contain the same frequencies, but that an adequate re-use factor is employed such as the three to one frequency re-use pattern illustrated in FIG. 6. When uplink FDMA channels are associated with corresponding downlink TDMA timeslots, the use of a three-to-one time re-use pattern on the downlink as disclosed above automatically gives rise to a three-to-one frequency re-use pattern on the uplink, thus achieving separation of signals. On the other hand, a one-to-one re-use frequency pattern can be achieved for the uplink using the configuration of FIG. 10 particularly when antenna 400 is a phased array.

The antenna 400, whether a multi-feed parabola or multi-element phased array, presents a number of RF ports containing a plurality of mobile uplink signals. A bank of low noise amplifiers 410 and downconverters 420 amplify these signals and coherently downconverts them using a common local oscillator 470 to a suitable intermediate frequency for amplification and filtering. The downconverted filtered and amplified signals are then applied to a bank of upconverters or transmitter modulators 430 which translate the signals to the C or Ka band while preserving their phase relationships before adding them in a combiner 440 and amplifying them in a traveling wave tube TWT power amplifier 450 for transmission to the HUB station through an antenna 460. It should be noted that the antenna 460 in FIG. 11 may be the same as the antenna 360 in FIG. 10, the C/Ka band receiver then being separated from the transmitter by means of a duplex filter. Moreover, both polarizations may be used in both directions in order to increase bandwidth utilization. Each polarization would then have associated with it half of the receiver bank 340 and half of the transmitter bank 430 connected to a separate traveling wave tube. Furthermore, a downlink antenna 300 and the uplink antenna 400 can also in principle be one and the same with the addition of transmit/receive duplexing filters for each beam, array element or sub-array, thus achieving double use of the same antenna aperture.

A description of the corresponding HUB station equipment may be found in the aforementioned U.S. patent application Ser. No. 08/179,953, entitled "A Cellular/Satellite Communication System With Improved Frequency Re-use", which is hereby incorporated by reference.

It will be appreciated by those of ordinary skill in the art that the number of timeslots, frequency bands and applications mentioned above are primarily for the purpose of illustration and are not meant to imply any limitation of the present invention. The present application contemplates any and all modifications that fall within the spirit and scope of the underlying invention disclosed and claimed herein.

What is claimed is:

1. A radio access method for facilitating communication between at least one first station and a plurality of second stations, comprising the steps of:

buffering at said first station each signal intended for transmission respectively to each of said second stations;

dividing said signals into equal length segments;

transmitting signal segments intended for a particular one of said second stations using a corresponding time slot in a regularly repeating time multiplex frame;

receiving at at least one of said second stations said signal segments transmitted by said first station in its corresponding time slot and assembling said signal segments from successive corresponding time slots to reconstruct said intended signal;

determining at said second station a transmit frequency channel uniquely associated with said corresponding receive time slot; and

buffering in said second station a signal intended for transmission to said first station and compressing it for transmission from said second station using said transmit frequency channel during substantially the entire time period that said second station is not receiving.

2. A radio access method for facilitating communication between a fixed station and a given number of mobile stations, comprising the steps of:

buffering at said fixed station signals to be transmitted to said mobile stations;

dividing said signals into segments;

transmitting the segments intended for a particular mobile station on time slots which are allocated for receiving at the particular mobile station, wherein said time slots regularly recur in a repetitive time multiplex frame which contains a number of time slots equal to said given number;

determining at each mobile station a transmit frequency channel uniquely corresponding to the mobile station's allocated time slot; and

buffering in at least one of said mobile stations a signal to be transmitted to said fixed station and transmitting the signal on said transmit frequency channel during more than one time slot in which said mobile station is not receiving.

3. A radio access method for facilitating communication between a first station and a predetermined number of second stations, comprising the steps of:

buffering in said first station signals to be transmitted to said second stations, said first and second stations having means for transmitting and receiving signals;

dividing said signals into segments;

transmitting the segments intended for a particular one of said second stations using only time slots that are

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- allocated to said particular second station for receiving, wherein said time slots regularly recur in a repetitive time multiplex frame containing a number of time slots equal to said predetermined number; and
- buffering in at least one of said second stations a signal to be transmitted to said first station and transmitting said signal during more than one time slot in which said second station is not receiving.
4. A radio access method according to claim 3, wherein said second stations have a receive period for approximately one-third of the time and a transmit period for approximately two-thirds of the time.
5. A radio access method according to claim 4, wherein said transmit period is composed of a first part in which said transmission occupies a first-half of said transmit channel frequency and a second part in which said transmission occupies a second-half of said transmit channel frequency.
6. A radio access method according to claim 4, wherein the transmit signal spectrum during said transmit period occupies a first-half of said transmit channel frequency spectrum during alternate transmit periods and a second-half of said transmit channel frequency during the other transmit periods.
7. A radio access method according to claim 3, wherein said second station transmissions that overlap in time are arranged so that they do not overlap in frequency during the time overlap.
8. A radio access method according to claim 7, wherein said non-overlap in frequency is arranged by division of a transmit frequency channel bandwidth into subbands and allocating a unique subband to each transmission overlapping in time with at least one other transmission.
9. A radio access method according to claim 7, wherein said non-overlap in frequency is arranged by frequency division of a transmit frequency channel bandwidth into a number of subbands one less than said predetermined number and each transmission occupies said subbands sequentially during an equal number of subdivisions of its transmit time period in a sequence that does not overlap with sequences used by other transmissions.
10. A radio access method according to claim 7, wherein said nonoverlap in frequency is arranged by division of a transmit frequency channel bandwidth into a number of subbands one less than said predetermined number and each transmission occupies a unique subband during each of its transmission time periods.
11. A radio access method according to claim 10, wherein said unique subband used by a particular transmission is varied from one transmit period to another in a nonoverlapping sequence relative to other transmissions.
12. A radio transmitter/receiver for communicating in both directions with a radio network, comprising:
- a timing control unit for sequencing transmit and receive functions;
 - an antenna switch for connecting an antenna alternately to the transmitter/receiver under the control of said timing control unit;
 - a receiver portion controlled by said timing control unit to receive a signal from said radio network during an allocated time slot in a time division multiplex frame period; and
 - a transmitter portion capable of being controlled by said timing control unit to transmit in a timeslot of a transmit time division multiplex frame having a different number of timeslots than in said receive time division multiplex frame period and selected such that

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- said transmit timeslot coincides with a portion of said receive time division multiplex frame period in which said receiver is not receiving.
13. A radio transmitter/receiver according to claim 12, wherein said timing control unit includes a microprocessor connector to an application specific integrated circuit.
14. A radio transmitter/receiver according to claim 12, wherein said timing control unit also controls the channel center frequency of said transmitter in dependence on said allocated received time slot.
15. A radio transmitter/receiver according to claim 12, wherein said receiver portion contains means for amplifying said received signal, filtering the amplifying received signals, down-converting the filtered received signal, sampling and analog digital converting of the down-converted received signal, buffering of said digitized received signal in a buffer memory and comprises digital signal processing means to process said samples in said buffer memory.
16. A radio transmitter/receiver according to claim 15, wherein said digital signal processing means determines time synchronization and frequency accuracy of said received signal, demodulates and error corrects the received signal, and processes said corrected received signal to obtain digital voice samples.
17. A radio transmitter/receiver according to claim 16, further including means to convert said digital voice samples to an acoustic voice sample.
18. A radio transmitter/receiver according to claim 12, wherein said transmitter portion comprises:
- means for converting an acoustic voice signal to digital voice samples;
 - digital signal processing means for processing said digital voice samples to obtain a digitally modulated signal for transmission;
 - up-converting means including a programmable frequency synthesizer for translating said digitally modulated signal to a final selected transmit channel frequency; and
 - transmit power amplifier means for amplifying said translated final frequency signal to a power level suitable for transmission compatible with the desired communications range.
19. A radio transmitter/receiver according to claim 18, wherein said up-converting means comprises a phase lock loop to translate digital modulation from a first modulated signal frequency to said final selected transmit channel frequency.
20. A radio transmitter/receiver according to claim 18, wherein said digital signal processing means operates to compress said digital voice samples to a reduced number of data symbols by means of a speech coding algorithm, encode said reduced number of data symbols using an error correction code, buffer said coded symbols in a memory so as to affect their transmission using only that portion of the time in which said receiver is not receiving, and modulate said encoded and buffered samples by conditioning them so as to represent a sequence of smoothly varying phases or amplitudes or both of a carrier frequency signal such that the spectrum of said modulated carrier signal is contained within a desired frequency region or subband.
21. A radio transmitter/receiver according to claim 20, wherein said desired frequency range is a subdivision or subband of a shared transmit frequency channel used by other transmitter/receivers in communication with said radio network station.
22. A radio transmitter/receiver according to claim 21, wherein operation in said subband is determined by said

digital signal processing means applying a phase ramp during said modulation conditioning equating to the desired frequency offset of said subband from the center of said shared channel.

23. A radio transmitter/receiver for communicating in both directions with a radio network, comprising:

- a timing control unit for sequencing transmit and receive functions repetitively within a given frame period;
- a receiver unit activated by said timing control unit to receive a signal from said radio network during a first allocated time slot with a length equal to a first portion of said frame period; and
- a transmitter unit capable of being activated by said timing control unit to transmit during a second allocated time slot with a length equal to a second portion of said frame period which does not overlap said first portion and in which said second time slot is substantially longer than said first time slot.

24. A radio transmitter/receiver according to claim 23, wherein said timing control unit includes a microprocessor and connected to application specific integrated circuit.

25. A radio transmitter/receiver according to claim 23, wherein said timing control unit also controls the channel center frequency of said transmitter in dependence on said allocated received time slot.

26. A radio transmitter/receiver according to claim 23, wherein said receiver portion contains means for amplifying said received signal, filtering said amplified signal, down-converting said filtered signal, sampling and analog-to-digital converting of the down-converted signal, buffering of said digitized signal in a buffer memory and digital signal processing means to process said samples in said buffer memory.

27. A radio transmitter/receiver according to claim 26, wherein said digital signal processing means determines time synchronization and frequency accuracy of said received signal, demodulates and error corrects the received signal, and processes said corrected received signal to obtain digital voice samples.

28. A radio transmitter/receiver according to claim 27, further including means to convert said digital voice samples to an acoustic voice signal.

29. A radio transmitter/receiver according to claim 23, wherein said transmitter portion comprises:

- means for converting an acoustic voice signal to digital voice samples;
- digital signal processing means for processing said digital voice samples to obtain a digitally modulated signal for transmission;
- up-converting means including a programmable frequency synthesizer to translate said digitally modulated signal to a final selected transmit channel frequency; and
- transmit power amplifier means for amplifying said translated final frequency signal to a power level suitable for transmission and compatible with the desired communications range.

30. A radio transmitter/receiver according to claim 29, wherein said up-converting means comprises a phase lock loop to translate digital modulation from a first modulated signal frequency to said final selected transmit channel frequency.

31. A radio transmitter/receiver according to claim 29, wherein said digital signal processing means operates to compress said digital voice samples into a reduced number of data symbols by means of a speech coding algorithm,

encode said reduced number of data symbols using an error correction code, buffer said encoded symbols in a memory so as to affect their transmission using only that portion of the time in which said receiver is not receiving, and modulate said encoded and buffered samples by conditioning them so as to represent a sequence of smoothly varying phases or amplitudes or both of a carrier frequency signal such that the spectrum of said modulated carrier signal is contained within a desired frequency region or subband.

32. A radio transmitter/receiver according to claim 31, wherein said desired frequency region is a subdivision or a subband of a shared transmit frequency channel used by other transmitter/receivers in communication with said radio network station.

33. A radio transmitter/receiver according to claim 32, wherein operation of said subband is determined by said digital signal processing means applying a phase ramp during said modulation conditioning equating to the desired frequency offset of said subband from the center of said shared channel.

34. A radio base station for communication with a plurality of outstations, comprising:

- multiplexing means for multiplexing digitized, compressed and error correction coded speech signals intended for said outstations into corresponding time slots in a TDM frame;
- modulating means to impress said multiplex digital signals on a transmitter carrier frequency signal;
- transmitter means coupled to antenna means to transmit said modulated carrier signals; and

a plurality of receiver means coupled to the antenna means for receiving signals from said outstations on a corresponding plurality of signal frequencies within a shared channel bandwidth, each receiver means receiving a signal from a particular outstation only when said transmitter means is not transmitting the time slot intended for reception at said particular outstation.

35. A radio base station according to claim 34, wherein said plurality of said receiver means is at least one less than said plurality of outstations.

36. A radio base station according to claim 34, in which said plurality of receiver means are coupled to digital signal processor means that sort the signals received from said plurality of receiver means in order to decode signals each having originated from different outstation transmitters.

37. A radio base station according to claim 36, wherein said sorting and decoding reconstructs a digitized voice signal transmitted from at least one of said outstations.

38. A radio base station according to claim 37, wherein said reconstructed, digitized voice signals from a plurality of outstations are remultiplexed for connection to a digital exchange employing time slot interchange.

39. A radio base station according to claim 34, wherein said multiplexing means is a digital exchange employing time slot interchange.

40. A radio base station according to claim 36, wherein signals received from one or more remote sites are also sorted and processed to provide improved reception of signals to said outstations.

41. A radio base station according to claim 34, wherein said plurality of receiver means is constructed by means for digitizing the composite of all received signals, processing said digitized composite signal with the aid of a digital filter bank to separate individual subband signals, sorting samples from said subband signals according to their originating outstations, and processing sorted samples corresponding to each outstation in order to reconstruct a digital voice signal transmitted by that outstation.

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42. A radio base station according to claim 41, wherein said processing performs the operations of demodulating the digitized composite signal, error correction decoding said demodulated signal and speech decoding said error corrected signal.

43. A radio base station according to claim 41, wherein said processing processes subband signals in order from strongest to weakest.

44. A radio base station according to claim 43, wherein an already processed stronger signal is then conditioned for subtraction from adjacent, weaker subband signals in order to enhance the isolation between adjacent subbands.

45. A radio base station according to claim 36, wherein said digital signal processor means jointly demodulates received signals originating from more than one of said outstations.

46. A method of communication for providing telephone communications between at least one outstation in a subscriber in a public switched telephone network comprising at least one orbiting satellite and at least one ground control station in communication with said satellite and said public switched telephone network, comprising the steps of:

transmitting from said satellite a signal bearing time division multiplex information comprised of at least one time slot in a repetitive TDMA frame period, each time slot being allocated for reception by one of said outstations;

receiving signal bursts from said satellite at said one outstation during its allocated time slot in said TDMA frame;

determining a transmit frequency channel uniquely associated with said allocated time slot; and

transmitting signal bursts using said transmit frequency channel from said one outstation to said satellite using substantially all of the remaining unused parts of the TDMA frame during which said outstation is not receiving.

47. A method for providing telephone communications between a number of outstations and a subscriber in a public switched telephone network comprised of at least one orbiting satellite and at least one ground control station in communication with said satellite and the public switched telephone network, comprising the steps of:

allocating to each of said outstations a time slot in a repeating time division multiplex frame, for reception from said satellite and an associated frequency channel for transmission to said satellite;

receiving signal bursts from said satellite at one of said outstations during its allocated time slot in said TDM frame; and

transmitting signal bursts using said allocated transmit frequency channel from said one of said outstations to said satellite during one or more of said time slots not allocated for reception at the same outstation.

48. A method for providing telephone communications between a first number of outstations and a subscriber in a

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public switch telephone network comprising at least one orbiting satellite and at least one ground control station in communication with said satellite and said public switch telephone network comprising the steps of:

allocating to each of said outstations a time slot in a repeating down-link time division multiplex frame period for reception from said satellite;

receiving signal bursts from said satellite at one of said outstations during said allocated time slot and said downlink TDM frame;

determining a second number of transmit frequency channels and a third number of transmit time slots in a repetitive uplink TDM frame to be used by said outstations for transmitting to said satellite, the product of said second and third numbers being equal to said first number;

allocating to each of said outstations a unique combination of one of said transmit frequency channels and transmit time slots depending on their respective allocated receive time slots such that the transmit time slot and receive time slot at any particular outstation do not overlap; and

transmitting signal bursts using said allocated transmit frequency channel in said allocated transmit time slot from at least one of said outstations to said satellite.

49. A method for providing telephone communication between a number of outstations and a subscriber in a public switched telephone network comprising at least one orbiting satellite and at least one ground control station in communication with said satellite and said public switched telephone network, comprising the steps of:

allocating to each of said outstations a time slot in a repeating down-link time division multiplex frame period for reception from said satellite;

receiving signal bursts from said satellite at one of said outstations during its allocated time slot and said downlink TDM frame;

determining a second number of transmit frequency channels and a third number of transmit codes from a set of code division multiple access codes to be used by said outstations for transmitting to said satellite, the product of said second and third numbers being equal to said first number;

allocating to each of said outstations a unique combination of one of said transmit frequency channels and one of said CDMA codes in dependence on their respective allocated receive time slots; and

transmitting signal bursts using said allocated transmit frequency channel and CDMA code from at least one of said outstations to said satellite using unused parts of said TDM frame period during which time said outstation is not receiving from said satellite.

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